

ESE 531: Digital Signal Processing

Lec 1: January 17, 2019
Introduction and Overview



Lecture Outline

- ❑ Course Topics Overview
- ❑ Learning Objectives
- ❑ Course Structure
- ❑ Course Policies
- ❑ Course Content
- ❑ What is DSP?
- ❑ DSP Examples



Course Topics Overview

- ❑ Discrete-Time (DT) Signals
- ❑ Time-Domain Analysis of DT Systems
- ❑ Discrete Fourier Transform (DFT)
- ❑ Fast Fourier Transform (FFT)
- ❑ Discrete-Time Fourier Transform (DTFT)
- ❑ z-Transform
- ❑ Sampling of Continuous Time Signals
- ❑ Data Converters and Modulation
- ❑ Upsampling/Downsampling
- ❑ Discrete-Time Filter Design



Learning Objectives

- ❑ Learn the fundamentals of digital signal processing
- ❑ Provide an understanding of discrete-time signals and systems and digital filters
- ❑ Enable you to apply DSP concepts to a wide range of fields
- ❑ Gain the ability to read the technical literature on DSP
- ❑ Apply the techniques learned in a final project encompassing many different application types



Learning Objectives

□ In other words...

□ Math, Math, Math*

*With MATLAB application for intuition



Course Structure

- ❑ TR Lecture, 4:30-6:00pm in DRLB A2
 - Start 5 minutes after, end 5 minutes early (~75-80min)
- ❑ Website (<http://www.seas.upenn.edu/~ese531/>)
 - Course calendar is used for all handouts (lectures slides, assignments, and readings)
 - Canvas used for assignment submission and grades
 - Piazza used for announcements and discussions



Course Structure

- ❑ Course Staff (complete info on course website)
- ❑ Instructor: Tania Khanna
 - Office hours – Wednesday 2-4 pm or by appointment
 - Email: taniak@seas.upenn.edu
 - Best way to reach me
- ❑ TAs:
 - Taishan Li
 - Office hours – WF 10am-11:30am in TBD
 - Mingxuan Sun
 - Office hours – TTh 3-4:30pm in TBD
- ❑ Grader: Yulai Weng



Course Structure

□ Lectures

- Statistically speaking, you will do better if you come to lecture
- Better if interactive, **everyone** engaged
 - Asking and answering questions
 - Actively thinking about material

□ Textbook

- A. V. Oppenheim and R. W. Schafer (with J. R. Buck), Discrete-Time Signal Processing. 3rd. Edition, Prentice-Hall, 2010
- Class will follow text structure... mostly



Course Structure - Assignments/Exams

- ❑ Homework – one week long (9 total)* [25%]
 - Due Sundays at midnight
 - Combination of book problems and matlab problems
 - Lowest grade dropped
- ❑ Project – two weeks long [30%]
 - Work in pairs
 - Combination of different DSP applications
- ❑ Midterm exam [20%]
- ❑ Final exam [25%]



Course Policies

See web page for full details

- ❑ Turn homework in Canvas
 - Anything handwritten/drawn must be clearly legible
 - Submit CAD generated figures, graphs, results when specified
 - NO LATE HOMEWORKS!
- ❑ Individual work (except project)
 - CAD drawings, simulations, analysis, writeups
 - May discuss strategies, but acknowledge help



Course Content

- ❑ Introduction
- ❑ Discrete Time Signals & Systems
- ❑ Discrete Time Fourier Transform
- ❑ Z-Transform
- ❑ Inverse Z-Transform
- ❑ Sampling of Continuous Time Signals
- ❑ Frequency Domain of Discrete Time Series
- ❑ Downsampling/Upsampling
- ❑ Data Converters, Sigma Delta Modulation
- ❑ Frequency Response of LTI Systems
- ❑ Signal Flow Representation
- ❑ Basic Structures for IIR and FIR Systems
- ❑ Design of IIR and FIR Filters
- ❑ Butterworth, Chebyshev, and Elliptic Filters
- ❑ Filter Banks
- ❑ Adaptive Filters
- ❑ Computation of the Discrete Fourier Transform
- ❑ Fast Fourier Transform

Course Content

ESE531 Spring 2019 Working Schedule

Wk	Lect.	Date	Lecture	Slides	Due	Reading
1	1	1/17 Th	Intro/Overview			review course webpage completely
2	2	1/22 T	Discrete Time Signals & Systems, Part 1			2.1-2.2
	3	1/24 Th	Discrete Time Signals & Systems, Part 2		HW 0	2.3-2.4
3	4	1/29 T	Discrete Time Fourier Transform			2.5-2.7
	5	1/31 Th	Z-Transform			3.0-3.1
		2/3 Su			HW 1	
4	6	2/5 T	Inverse Z-Transform			3.3
	7	2/7 Th	Sampling and Reconstruction			4.0-4.3
		2/10 Su			HW 2	
5	8	2/12 T	DT/CT Processing of CT/DT Signals			4.3-4.4.1
	9	2/14 Th	DT/CT Processing of CT/DT Signals (con't), Impulse Invariance			4.3-4.4.1
		2/17 Su			HW 3	
6	10	2/19 T	Downsampling/Upsampling and Practical Interpolation			4.4.2-4.6.3
	11	2/21 Th	Non-Integer and Multi-rate Sampling			4.6.4-4.7
		2/24 Su			HW 4	
7	12	2/26 T	Data Converters and Noise Shaping			4.8-4.9
	13	2/28 Th	Frequency Response of LTI Systems			5.0-5.3
		3/3 Su			HW 5	
8		3/5 T	SPRING BREAK -- no class			
		3/7 Th	SPRING BREAK -- no class			
9		3/12 T	Midterm Exam, during class in TBD			
10	14	3/14 Th	All-pass Systems, Min Phase Decomposition			5.4-5.6
	15	3/19 T	Generalized Linear Phase Systems			5.7
	16	3/21 Th	Design of IIR Filters			7.0-7.2

What is DSP

Bohemian Rhapsody



❑ <https://www.youtube.com/watch?v=l3iIJu08t-Y>

Bohemian Rhapsody



❑ <https://www.youtube.com/watch?v=-yE8SYzZ6Eo>



DSP is Everywhere

- ❑ Sound applications
 - Compression, enhancement, special effects, synthesis, recognition, echo cancellation,...
 - Cell phones, MP3 players, movies, dictation, text-to-speech,...
- ❑ Communication
 - Modulation, coding, detection, equalization, echo cancellation,...
 - Cell Phones, dial-up modem, DSL modem, Satellite Receiver,...
- ❑ Automotive
 - ABS, GPS, Active Noise Cancellation, Cruise Control, Parking,...



DSP is Everywhere (con't)

- ❑ Medical
 - Magnetic Resonance, Tomography, Electrocardiogram, Biometric Monitoring...
- ❑ Military
 - Radar, Sonar, Space photographs, remote sensing,...
- ❑ Image and Video Applications
 - DVD, JPEG, Movie special effects, video conferencing...
- ❑ Mechanical
 - Motor control, process control, oil and mineral prospecting,...



Signal Processing

- ❑ Humans are the most advanced signal processors
 - speech and pattern recognition, speech synthesis,...
- ❑ We encounter many types of signals in various applications
 - Electrical signals: voltage, current, magnetic and electric fields,...
 - Mechanical signals: velocity, force, displacement,...
 - Acoustic signals: sound, vibration,...
 - Other signals: pressure, temperature, biometrics...
- ❑ Most real-world signals are analog
 - They are continuous in time and amplitude
 - Convert to voltage or currents using sensors and transducers



Signal Processing (con't)

- ❑ Analog circuits process these signals using
 - Resistors, Capacitors, Inductors, Amplifiers,...
- ❑ Analog signal processing examples
 - Audio processing in FM radios
 - Video processing in traditional TV sets

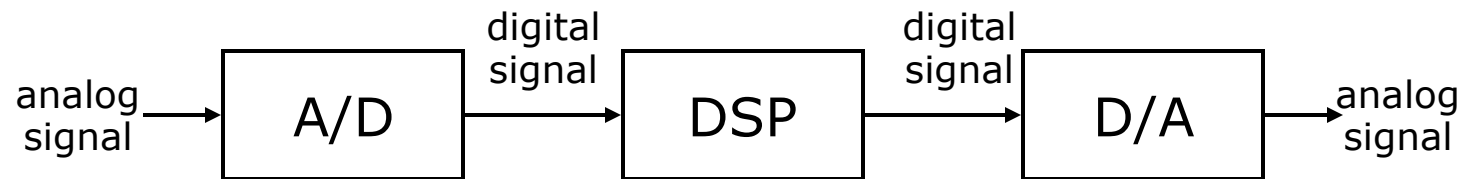


Limitations of Analog Signal Processing

- ❑ Accuracy limitations due to
 - Component tolerances
 - Undesired nonlinearities
- ❑ Limited repeatability due to
 - Tolerances
 - Changes in environmental conditions
 - Temperature
 - Vibration
- ❑ Sensitivity to electrical noise
- ❑ Limited dynamic range for voltage and currents
- ❑ Inflexibility to changes
- ❑ Difficulty of implementing certain operations
 - Nonlinear operations
 - Time-varying operations
- ❑ Difficulty of storing information

Digital Signal Processing

- ❑ Represent signals by a sequence of numbers
 - Sampling and quantization (or analog-to-digital conversion)
- ❑ Perform processing on these numbers with a digital processor
 - Digital signal processing
- ❑ Reconstruct analog signal from processed numbers
 - Reconstruction or digital-to-analog conversion



- Analog input → analog output
 - Eg. Digital recording music
- Analog input → digital output
 - Eg. Touch tone phone dialing, speech to text
- Digital input → analog output
 - Eg. Text to speech
- Digital input → digital output
 - Eg. Compression of a file on computer



Pros and Cons of Digital Signal Processing

❑ Pros

- Accuracy can be controlled by choosing word length
- Repeatable
- Sensitivity to electrical noise is minimal
- Dynamic range can be controlled using floating point numbers
- Flexibility can be achieved with software implementations
- Non-linear and time-varying operations are easier to implement
- Digital storage is cheap
- Digital information can be encrypted for security
- Price/performance and reduced time-to-market

❑ Cons

- Sampling causes loss of information
- A/D and D/A requires mixed-signal hardware
- Limited speed of processors
- Quantization and round-off errors

DSP Examples



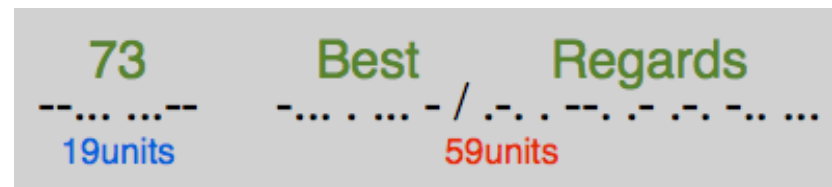
Example I: Audio Compression

- ❑ Compress audio by 10x without perceptual loss of quality
- ❑ Sophisticated processing based on models of human perception
- ❑ 3MB files instead of 30MB
 - Entire industry changed in less than 10 years!

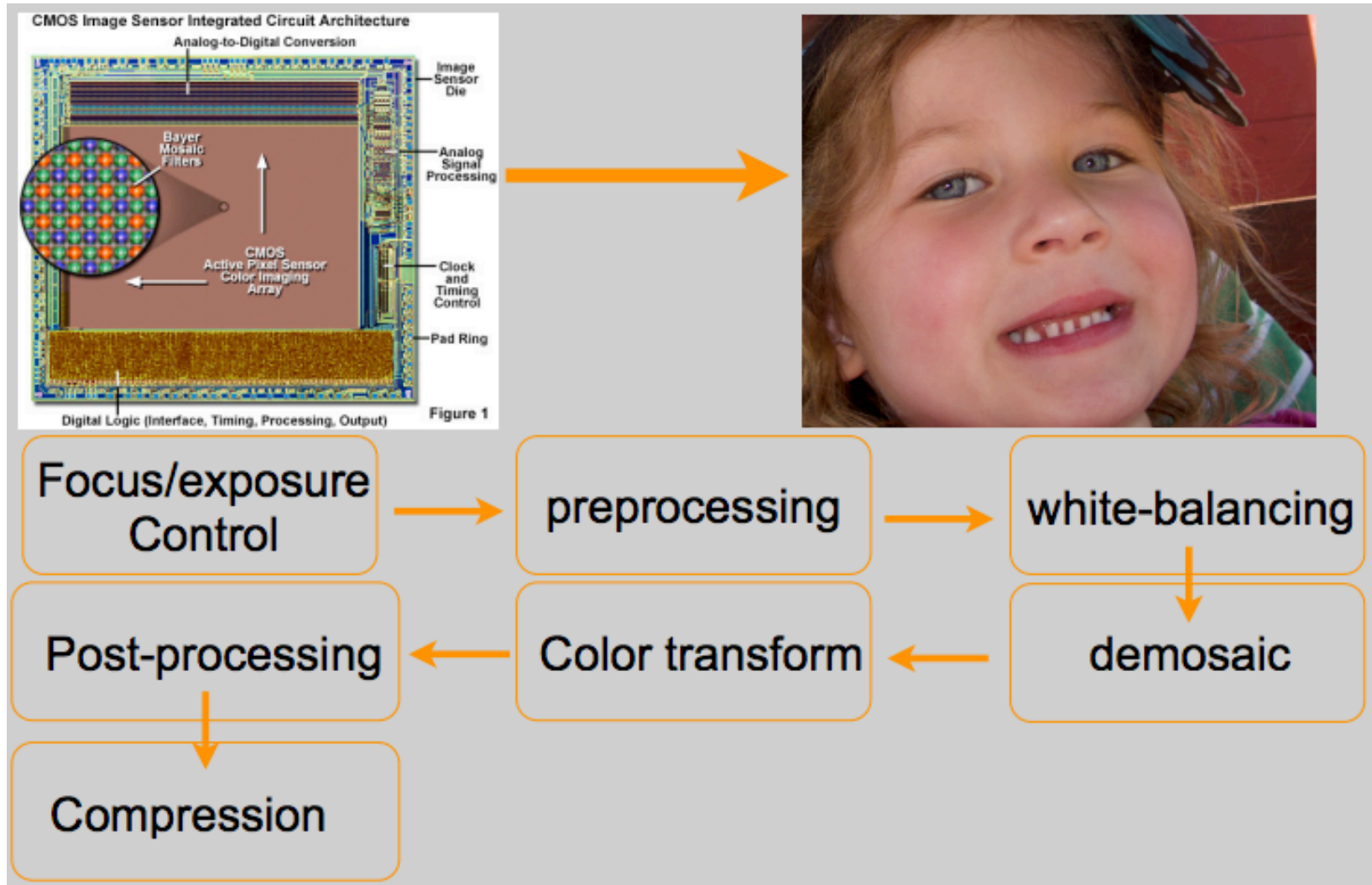
Historical Forms of Compression

- ❑ Morse code: dots (1 unit) dashes (3 units)
 - Code Length inversely proportional to frequency of character
 - E (12.7%) = . (1 unit) Q (0.1%) = --.- (10 units)

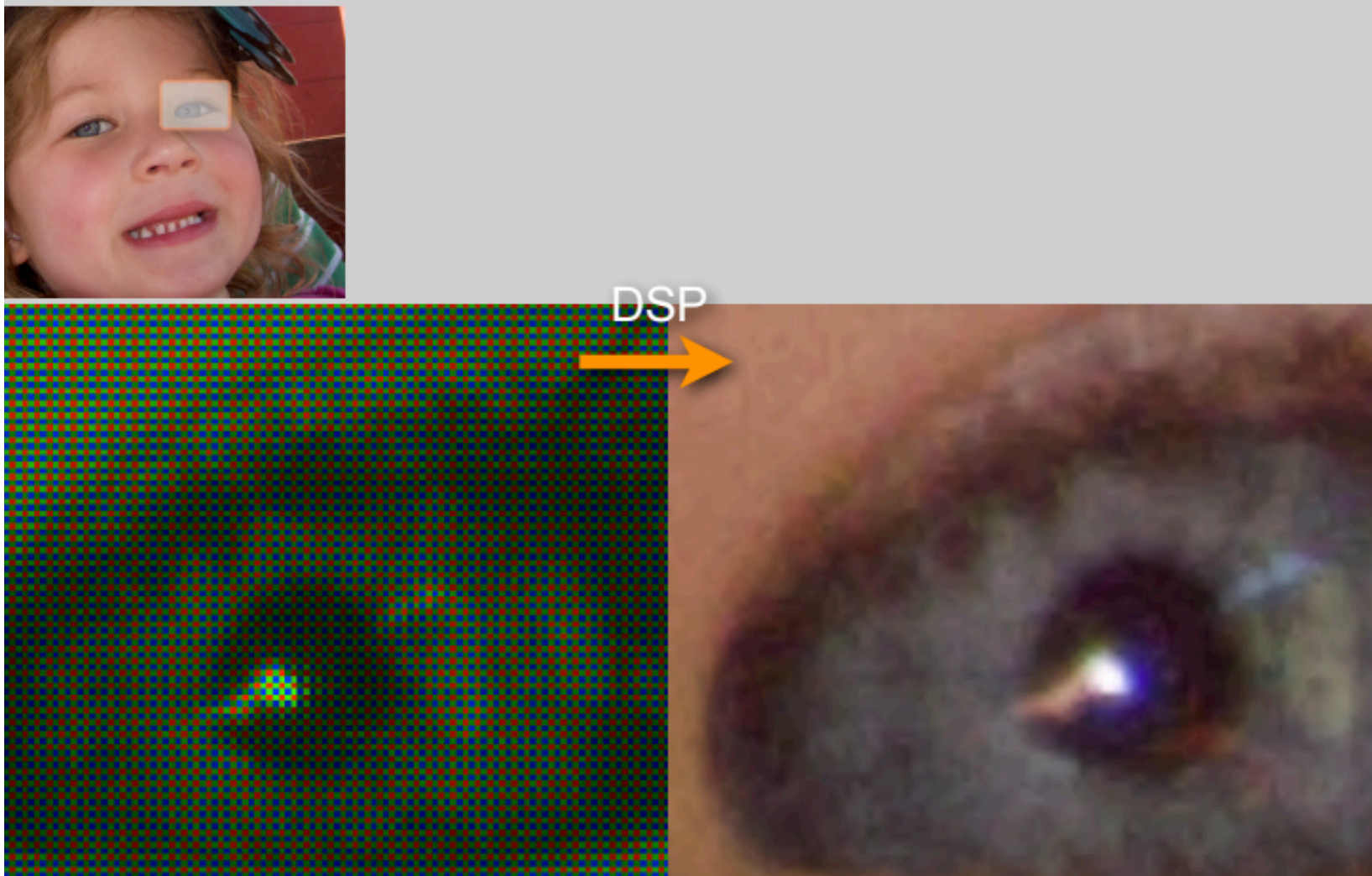
- ❑ “92 Code”
 - Used by Western-Union in 1859 to reduce BW on telegraph lines by numerical codes for frequently used phrases
 - 1 = wait a minute
 - 73 = Best Regards
 - 88 = Loves and Kisses



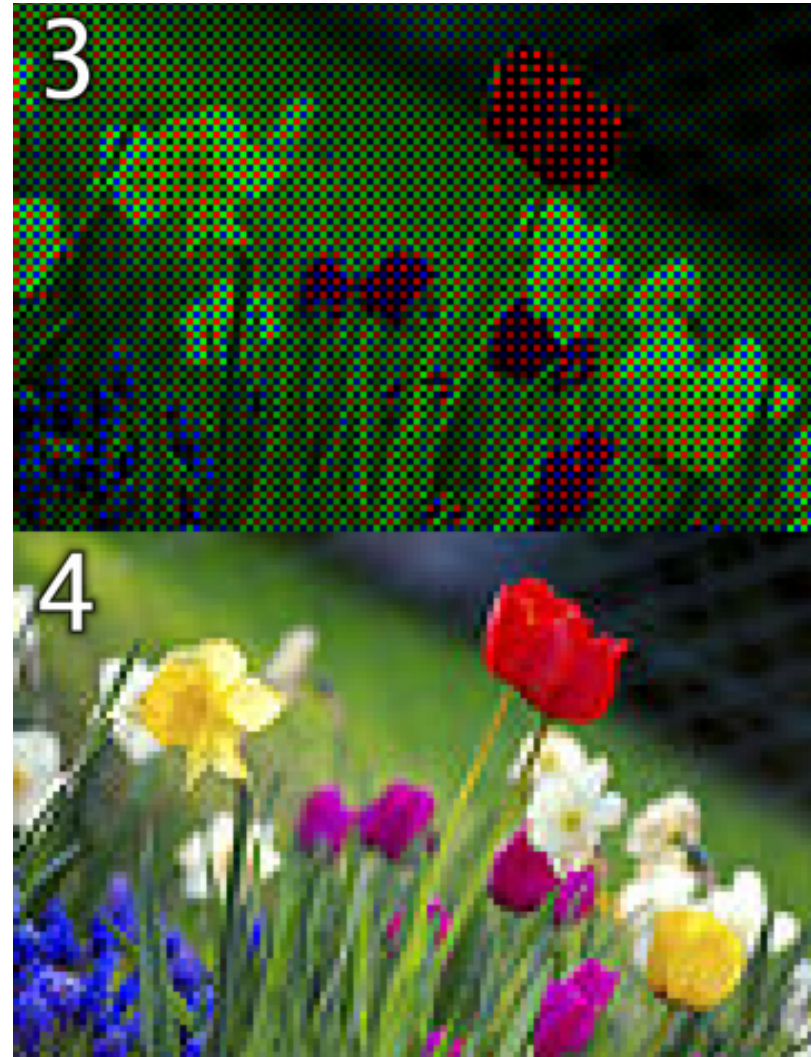
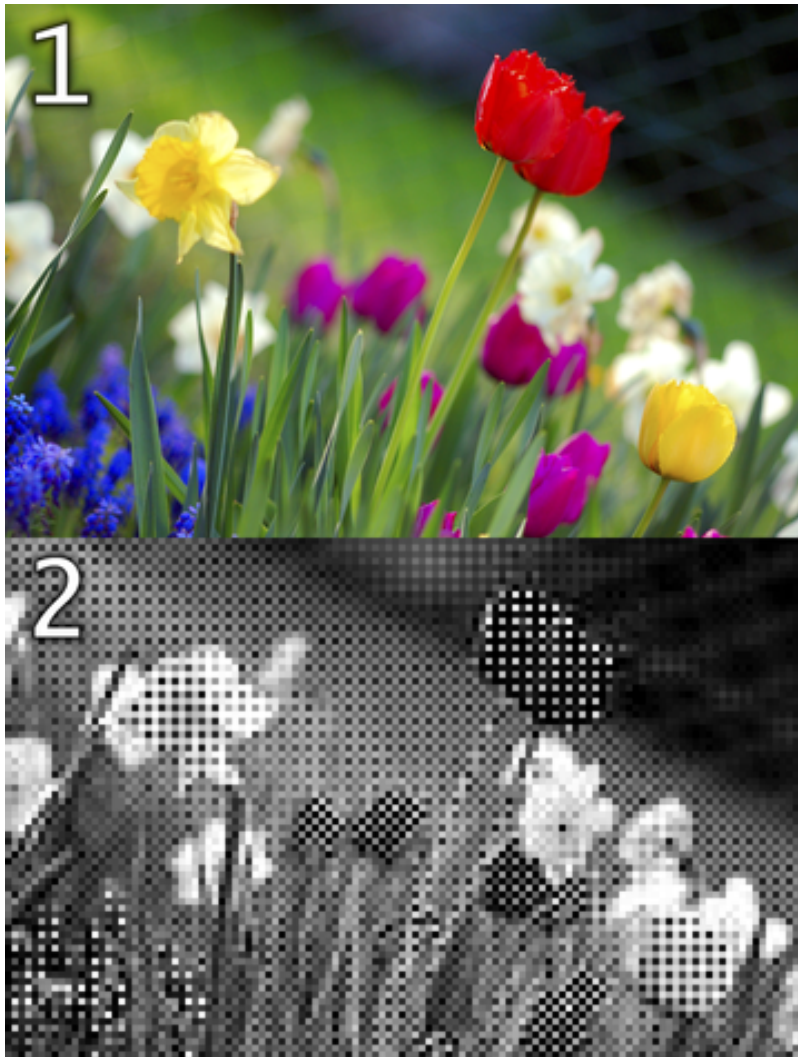
Example II: Digital Imaging Camera



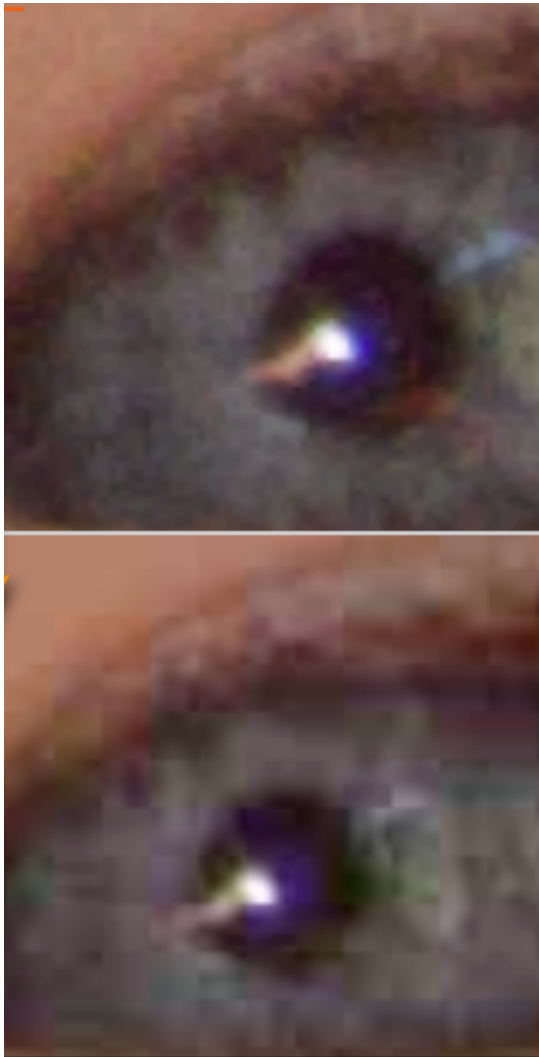
Example II: Digital Imaging Camera



Example II: Digital Imaging Camera



Example II: Digital Imaging Camera



- ❑ Compression of 40x without perceptual loss of quality.
- ❑ Example of slight overcompression: difference enables 60x compression!

Computational Photography

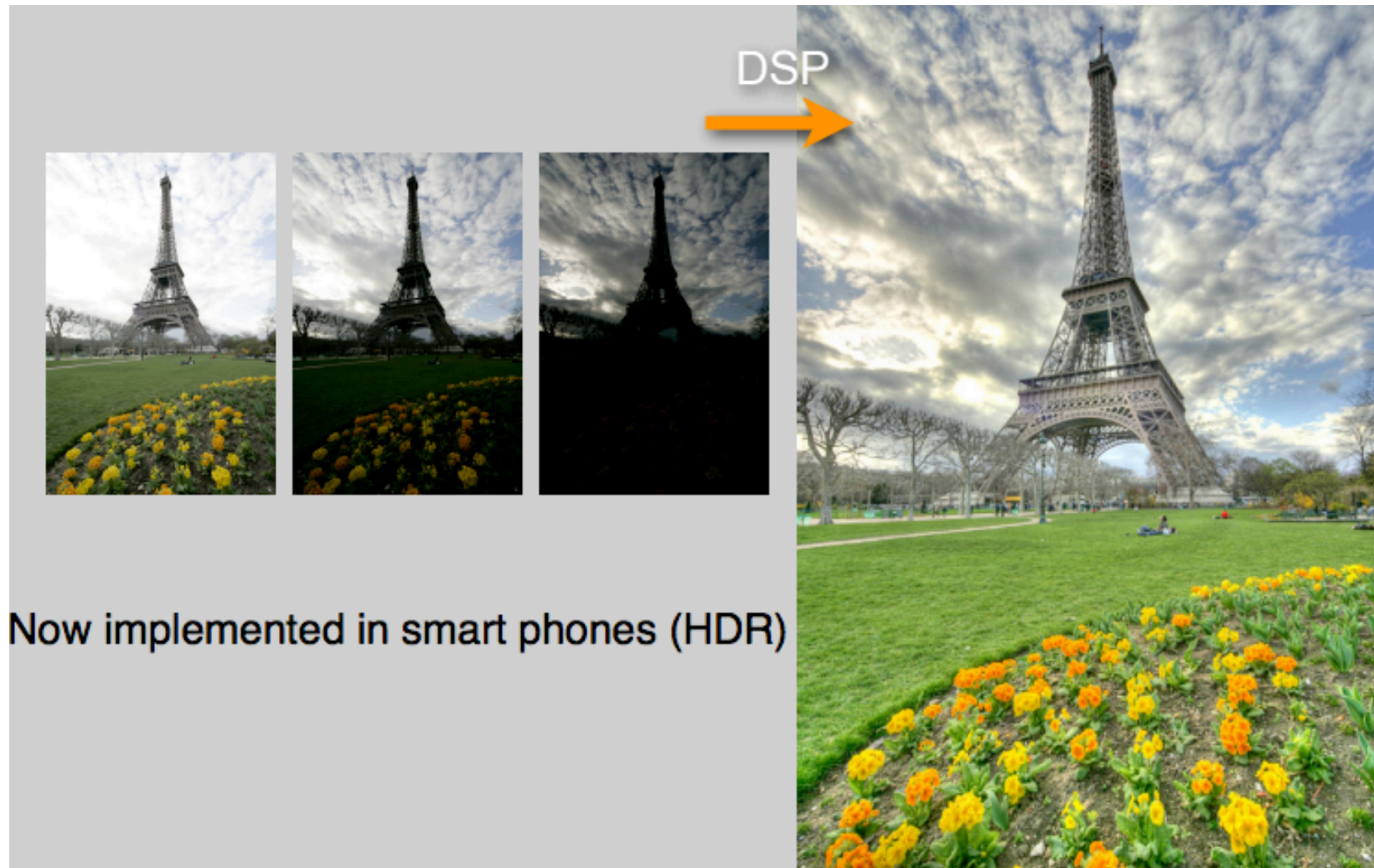


Image Processing

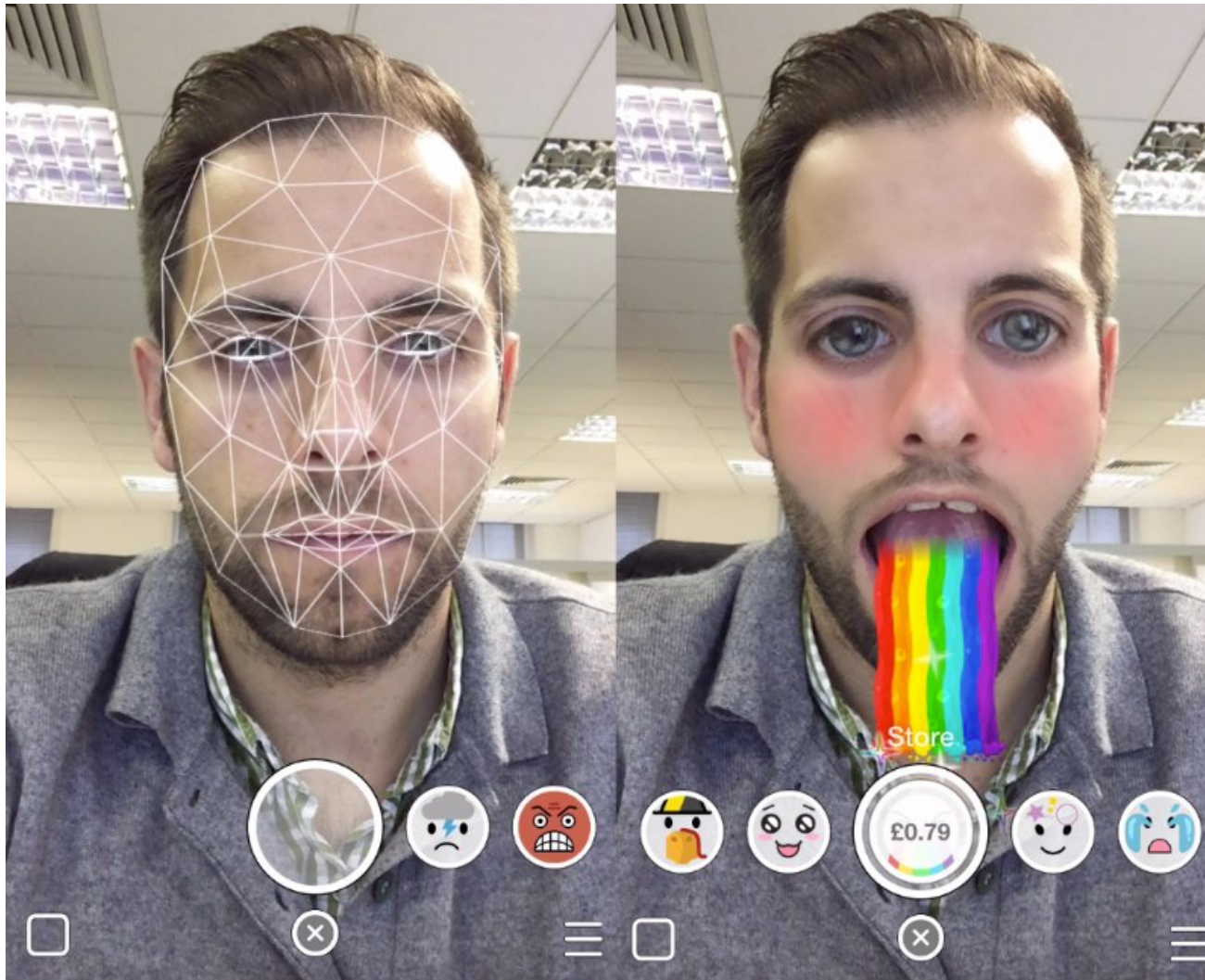


Image Processing - Saves Lives

Canadian 'swirl face'
Thailand

jailed in

August 15, 2008

☆ Rea

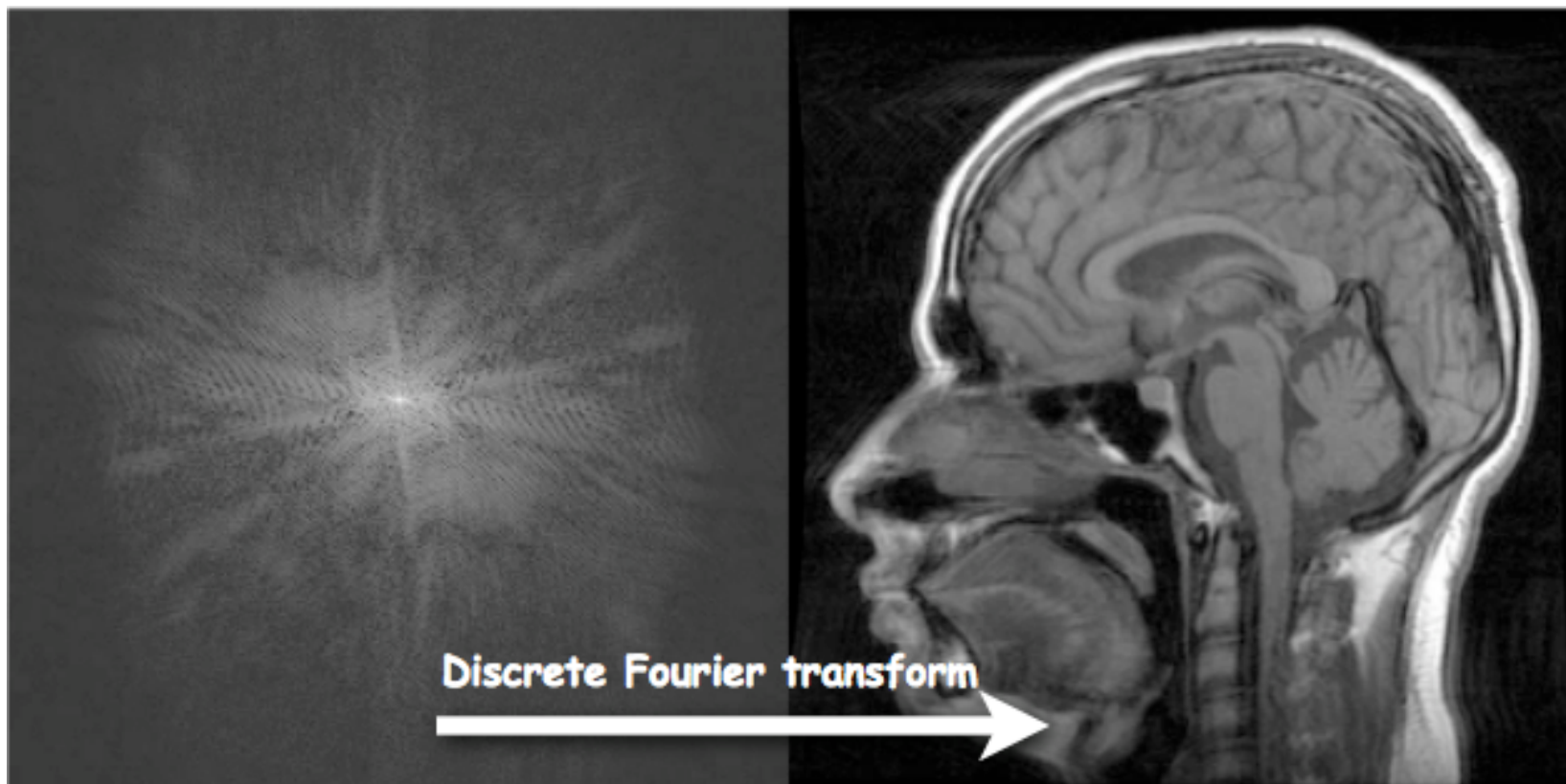


Images released by Interpol in 2007 show the 'unswirling' of the internet pictures that led to the capture of Christopher Paul Neil.

Example III: MRI

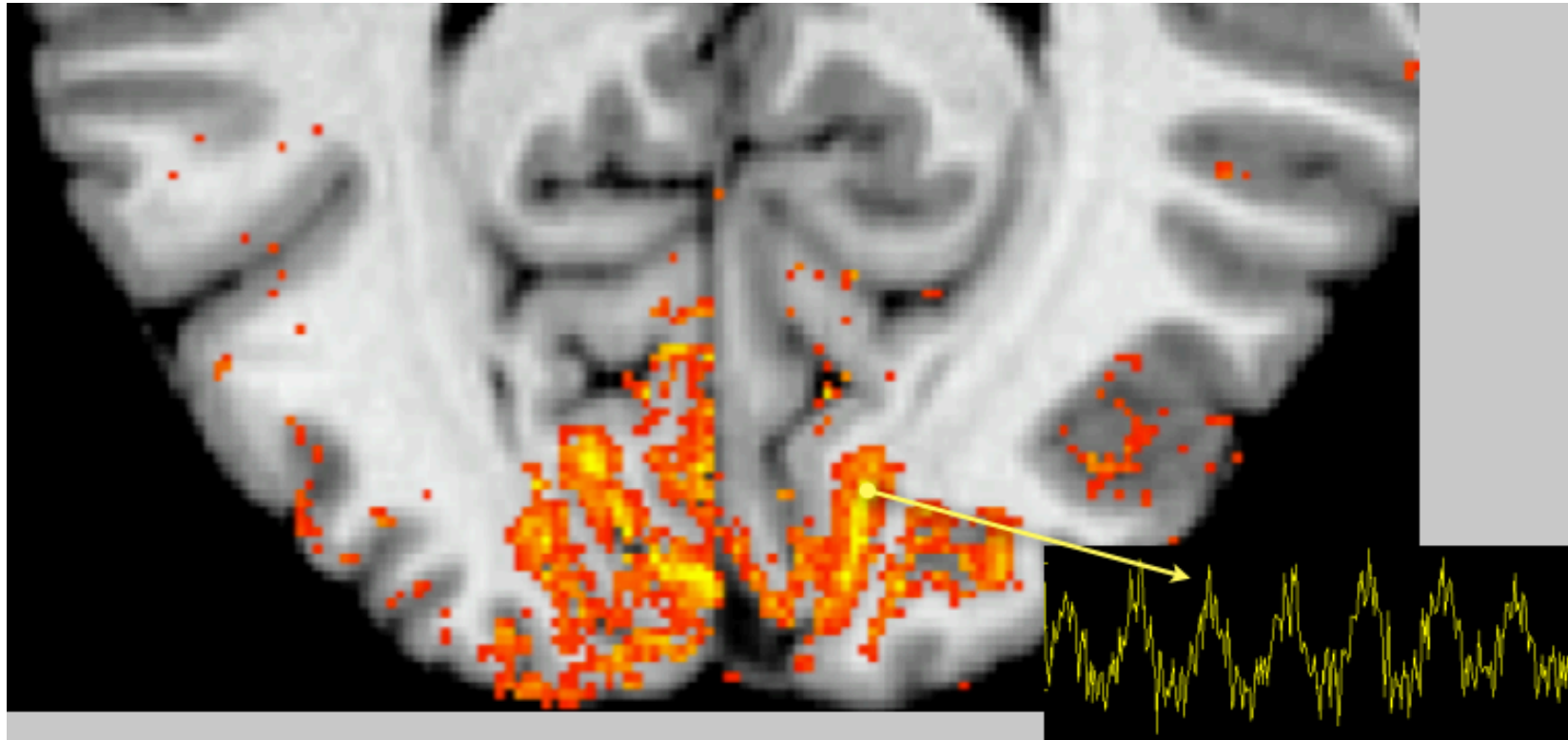
k-space (raw data)

Image



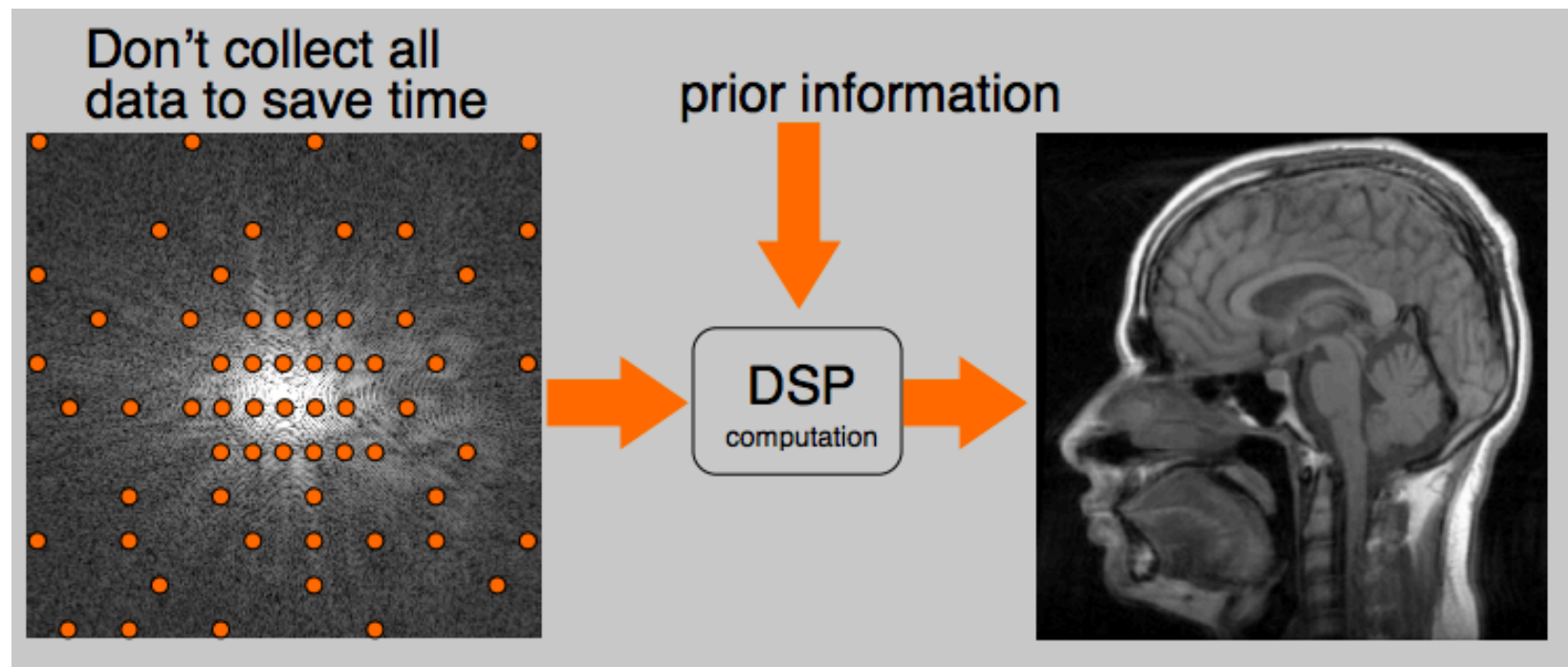
fMRI example

- Sensitivity to blood oxygenation
 - response to brain activity Convert from one signal to another



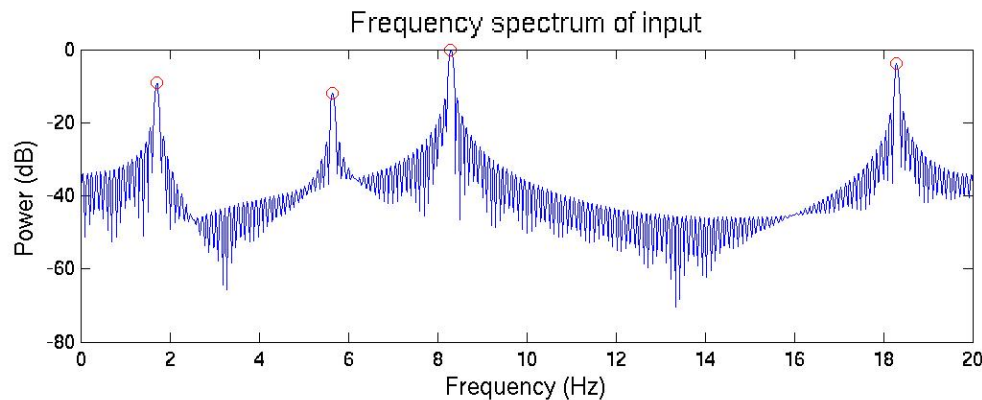
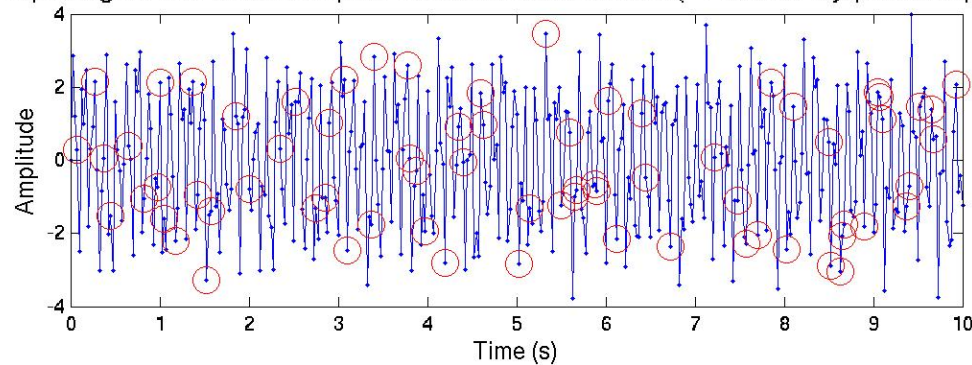
Compressive Sampling

- ❑ Compression meets sampling



Example: Sum of Sinusoids

Input signal with undersampled measurements circled ($\sim 17.5\%$ of Nyquist samples)



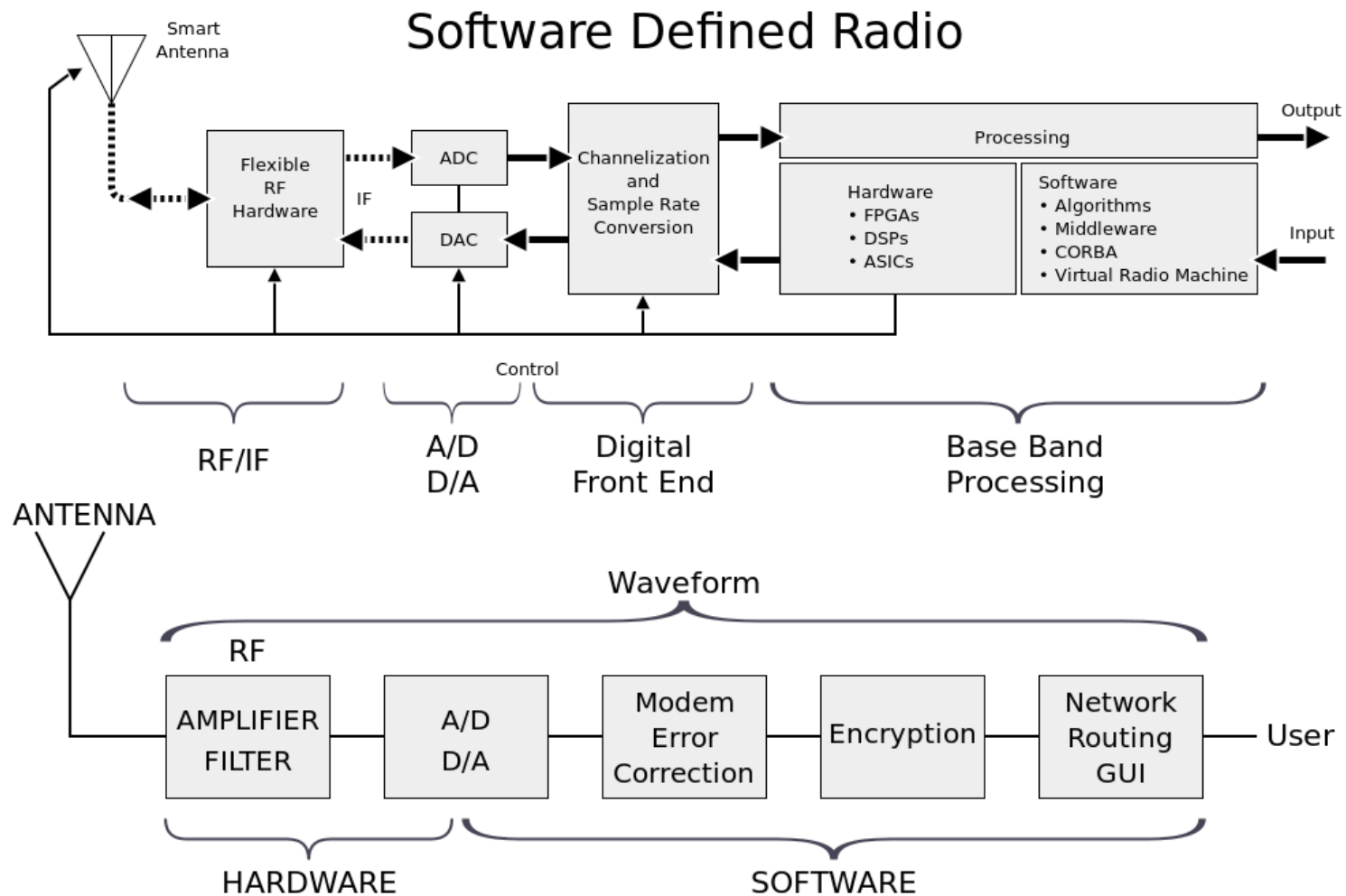
- Sense signal randomly M times
 - $M > C \cdot \mu^2(\Phi, \Psi) \cdot S \cdot \log N$
- Recover with linear program



Example IV: Software Defined Radio

- ❑ Traditional radio:
 - Hardware receiver/mixers/demodulators/filtering
 - Outputs analog signals or digital bits
- ❑ Software Defined Radio:
 - Uses RF front end for baseband signal
 - High speed ADC digitizes samples
 - All processing chain done in software

Software Defined Radio





Software Defined Radio

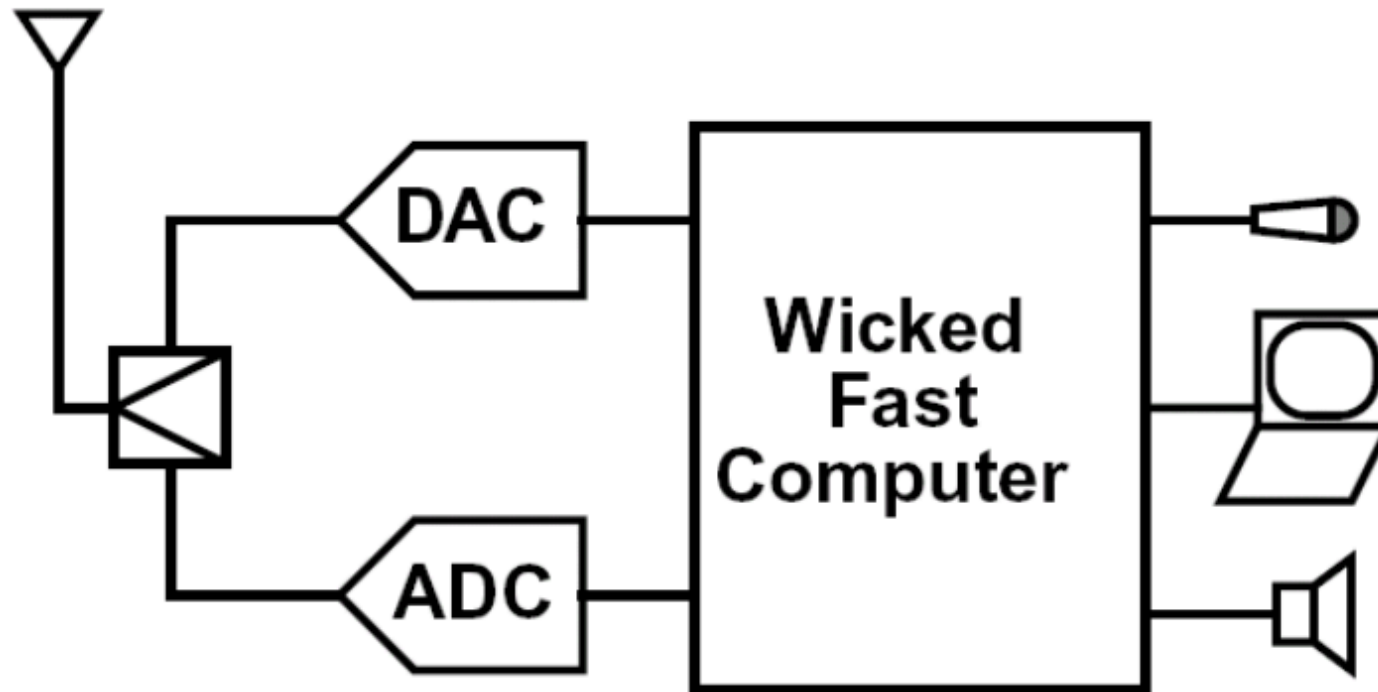
- ❑ Advantages:

- Flexibility
- Upgradable
- Sophisticated processing
- Ideal Processing chain
- not approximate like in analog hardware

- ❑ Already used in consumer electronics

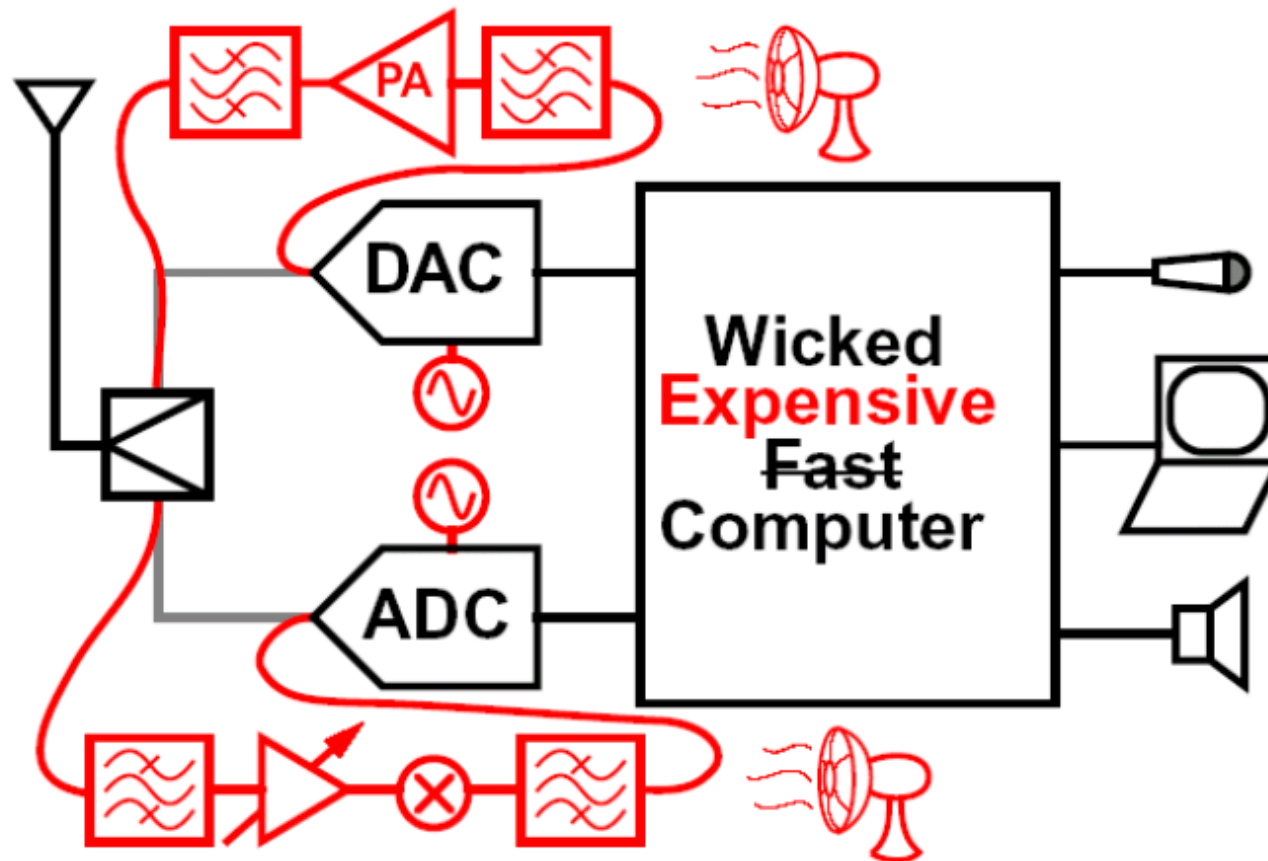
- Cellphone baseband processors
- Wifi, GPS, etc....

Software Radio Vision



[Schreier, "ADCs and DACs: Marching Towards the Antenna," GIRAFE workshop, ISSCC 2003]

Software Radio Reality



[Schreier, "ADCs and DACs: Marching Towards the Antenna," GIRAFE workshop, ISSCC 2003]



Shameless Plug

- ❑ If you are interested in how Analog to digital converters work and how to make them
- ❑ Take ESE 568!
- ❑ Good to know both sides of the system



Future of ADC design

- ❑ Today's ADCs are extremely well optimized
- ❑ For non-incremental improvements, we must explore new ideas in signal processing that tackle ADC inefficiency at the system level
 - Compressed sensing
 - Finite innovation rate sampling
 - Other ideas?

Filter Design Example





Optimal Filter Design

- ❑ Window method
 - Design Filters heuristically using windowed sinc functions
- ❑ Optimal design
 - Design a filter $h[n]$ with $H(e^{j\omega})$
 - Approximate $H_d(e^{j\omega})$ with some optimality criteria - or satisfies specs.

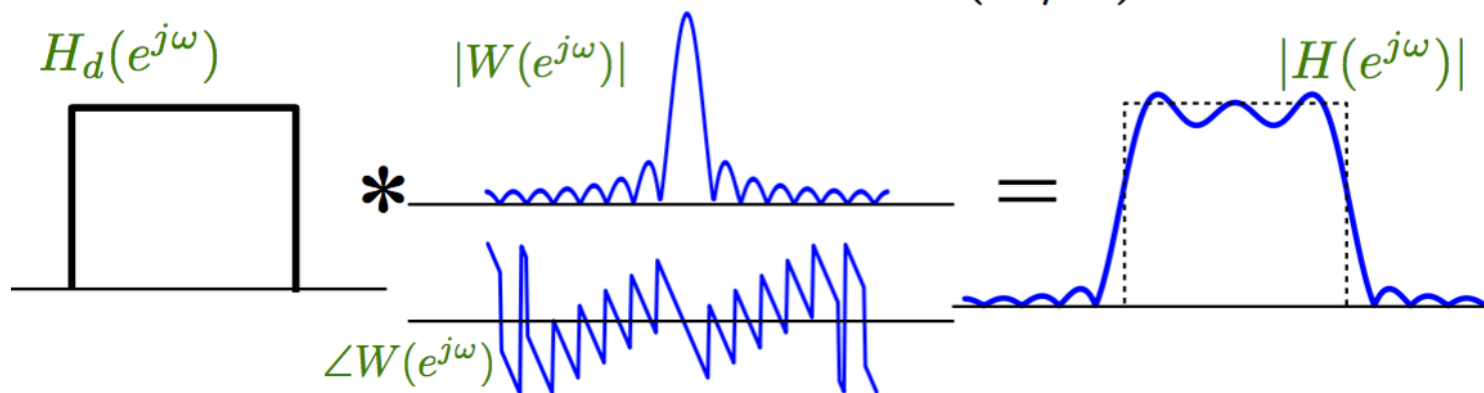
FIR Design by Windowing

- Desired filter,

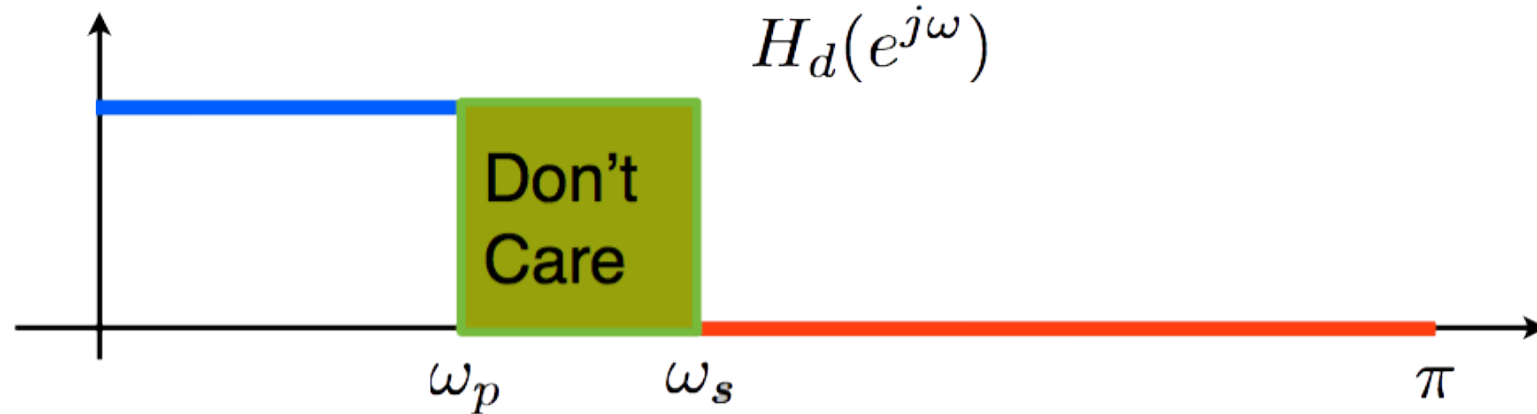
$$H(e^{j\omega}) = H_d(e^{j\omega}) * W(e^{j\omega})$$

- For Boxcar (rectangular) window

$$W(e^{j\omega}) = e^{-j\omega \frac{M}{2}} \frac{\sin(\omega(M+1)/2)}{\sin(\omega/2)}$$



FIR Design by Optimality



- Least Squares:

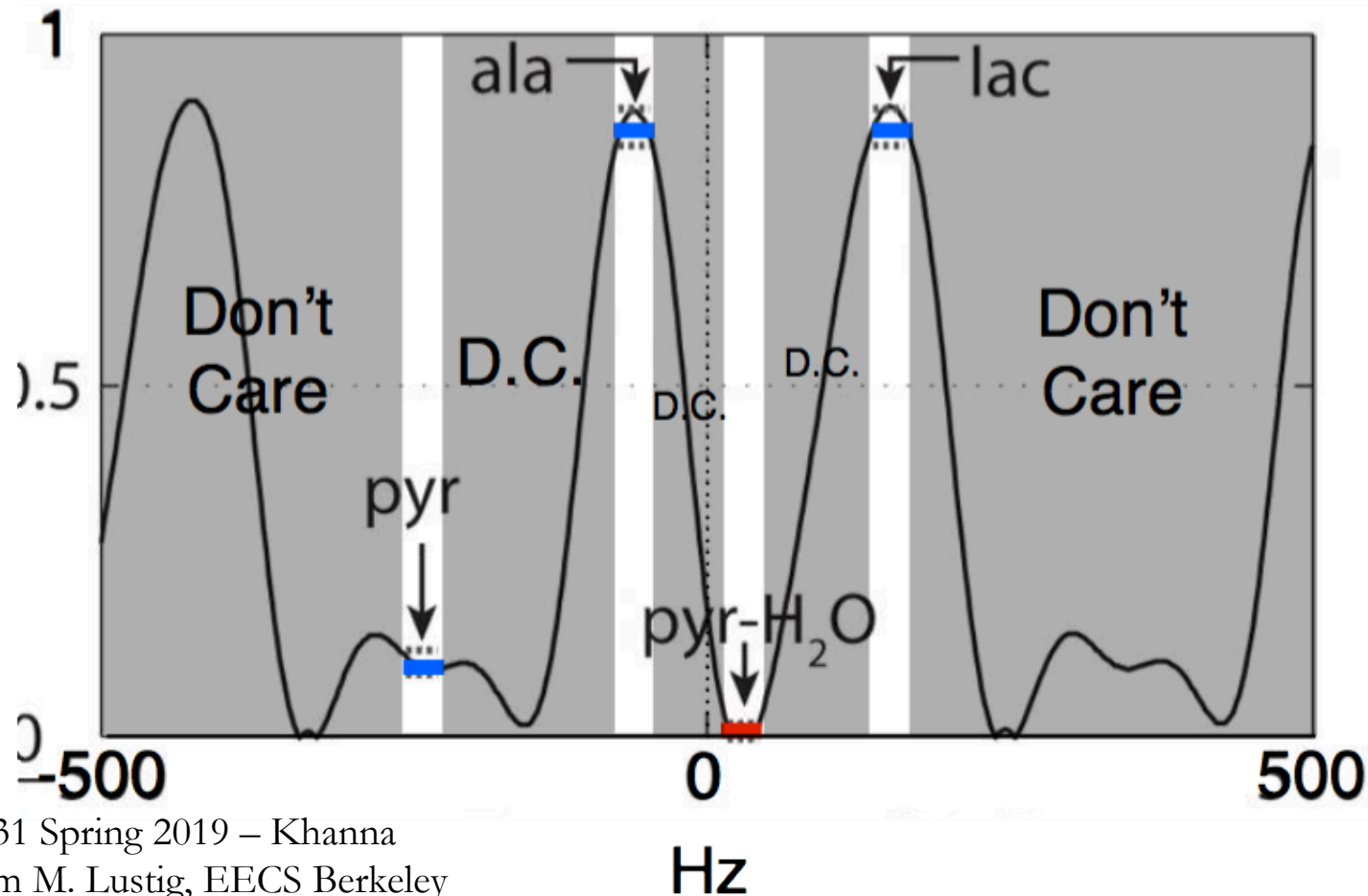
$$\text{minimize} \int_{\omega \in \text{care}} |H(e^{j\omega}) - H_d(e^{j\omega})|^2 d\omega$$

- Variation: Weighted Least Squares:

$$\text{minimize} \int_{-\pi}^{\pi} W(\omega) |H(e^{j\omega}) - H_d(e^{j\omega})|^2 d\omega$$

Example of Complex Filter

- ❑ Larson et. al, “Multiband Excitation Pulses for Hyperpolarized ^{13}C Dynamic Chemical Shift Imaging” JMR 2008;194(1):121-127
- ❑ Need to design 11 taps filter with following frequency response:





Admin

- ❑ Find web, get text, start HW 0 and assigned reading...
 - <http://www.seas.upenn.edu/~ese531>
 - <https://piazza.com/upenn/spring2019/ese531/>
 - <https://canvas.upenn.edu/>
- ❑ Remaining Questions?